## IN THE CLAIMS

1. (original) A method for separating a plurality of mixed signals into a plurality of component signals comprising the steps of:

- (a) producing a plurality of discrete Fourier transform(DFT) values corresponding to frequency components of an input segment of said mixed signals;
- (b) updating cross correlation estimation matrices using said DFT values;
- (c) computing, using a cost-function minimization process, an update value for a plurality of filter coefficients for a finite impulse response (FIR) filter using said cross correlation estimation values;
- (d) filtering said mixed signals using said FIR filter having said updated filter coefficients to separate said mixed signals into one or more component signals; and
- (e) iteratively repeating steps (a), (b), (c) and (d) for successive input segments of saidmixed signal.
- 2. (original) The method of claim 1 wherein step (c) further comprises the substeps of:
- (c1) transforming said update filter coefficients from the frequency domain into the time domain;

(c2) zeroing any filter coefficients having a value other than zero for any time that is greater than a predefined time Q, where Q is less than a value of an input-segment length, T; thereby producing a set of constrained time domain filter coefficients; and

- (c3) transforming said adjusted time domain filter coefficients from the time domain back into the frequency domain.
- (original) The method of claim 1 wherein said costfunction minimization process is a gradient descent process.
- 4. (original) The method of claim 1 wherein said computation of said filter coefficient update values further includes an adaptation of update step sizes based on a normalization factor.
- 5. (currently amended) The method of claim 4 wherein said update values, including said update step-size adaptation, are computed according to

$$\Delta_{\iota} \mathbf{W}(\omega) = -\frac{\mu}{h} \nabla_{w} E$$

where  $\mu$  is a fixed learning constant, h is a weighting factor for the step-size adaptation, and E is a filter-parameter cost function operating on a square difference respecting a diagonal covariance of said component signals, and E is a gradient step for E.

6. (original) The method of claim 1 wherein a running average of said cross correlation estimation values are produced according to

$$\hat{\mathbf{R}}_{x}(t,\omega) = (1-\gamma)\hat{\mathbf{R}}_{x}(t,\omega) + \gamma \mathbf{x}(t,\omega)\mathbf{x}^{H}(t,\omega).$$

where  $x(t,\omega)$  is the mixed signal in the frequency domain.

7. (original) An apparatus for separating a plurality of mixed signals into a plurality of component signals comprising:

means for producing a plurality of discrete Fourier transform (DFT) values corresponding to frequency components of an input segment of said mixed signals;

means for updating cross correlation estimation matrices using said DFT values;

a cost-function minimization processor for computing an update value for a plurality of filter coefficients for a finite impulse response (FIR) filter using said cross correlation estimation values; and

an FIR filter having said updated filter coefficients to separate said mixed signals into one or more component signals.

- 8. (original) The apparatus of claim 7 wherein said costfunction minimization processor further comprises:
- a first transformer for transforming said update filter coefficients from the frequency domain into the time domain;

means for zeroing any filter coefficients having a value other than zero for any time that is greater than a predefined time Q, where Q is less than a value of an input-segment length, T; thereby producing a set of constrained time domain filter coefficients; and

a second transformer for transforming said adjusted time domain filter coefficients from the time domain back into the frequency domain.

- 9. (original) The apparatus of claim 8 wherein said first transformer uses an inverse Fourier transform and said second transformer uses a Fourier transform.
- 10. (currently amended) The apparatus of claim 7 wherein said computation of said-filter-coefficient update values

  further includes an adaptation of update step-sizes-based on a normalization factor cost-function minimization processor carries out a gradient descent process.
- 11. (original) The apparatus of claim 7 wherein said computation of said filter coefficient update values further includes an adaptation of update step sizes based on a normalization factor.
- 12. (currently amended) The apparatus of claim 11 wherein said update values, including said update step-size adaptation, are computed according to

$$\Delta_{t} \mathbf{W}(\omega) = -\frac{\mu}{h} \nabla_{w} E$$

where  $\mu$  is a fixed learning constant, h is a weighting factor for the step-size adaptation, and E is a filter-parameter cost function operating on a square difference respecting a diagonal covariance of said component signals, and E is a gradient step for E.

13. (original) The apparatus of claim 7 wherein a running average of said cross correlation estimation values are produced according to

$$\hat{\mathbf{R}}_{x}(t,\omega) = (1-\gamma)\hat{\mathbf{R}}_{x}(t,\omega) + \gamma \mathbf{x}(t,\omega)\mathbf{x}^{H}(t,\omega).$$

where  $x(t,\omega)$  is the mixed signal in the frequency domain.

- 14. (original) The apparatus of claim 7 further comprising a voice recognition system for processing at least one of said component signals.
- 15. (original) A computer readable storage medium containing a program that, when executed upon a general purpose computer system, causes said general purpose computer system to become a specific purpose computer system that performs a method for separating a plurality of mixed signals into a plurality of component signals comprising the steps of:
- (a) producing a plurality of discrete Fourier transform(DFT) values corresponding to frequency components of an input segment of said mixed signals;

(b) updating cross correlation estimation matrices using said DFT values;

- (c) computing, using a cost-function minimization process, an update value for a plurality of filter coefficients for a finite impulse response (FIR) filter using said cross correlation estimation values;
- (d) filtering said mixed signals using said FIR filter having said updated filter coefficients to separate said mixed signals into one or more component signals; and
- (e) iteratively repeating steps (a), (b), (c) and (d) for successive input segments of said mixed signal.
- 16. (original) The computer readable medium of claim 15 wherein step (c) further comprises the substeps of:
- (c1) transforming said update filter coefficients from the frequency domain into the time domain;
- (c2) zeroing any filter coefficients having a value other than zero for any time that is greater than a predefined time Q, where Q is less than a value of an input-segment length, T; thereby producing a set of constrained time domain filter coefficients; and
- (c3) transforming said adjusted time domain filter coefficients from the time domain back into the frequency domain.

Docket No.: SAR 13666 (PATENT)

- 17. (original) The computer readable medium of claim 15 wherein said cost-function minimization process is a gradient descent process.
- 18. (original) The computer readable medium of claim 15 wherein said computation of said filter coefficient update values further includes an adaptation of update step sizes based on a normalization factor.
- 19. (original) The computer readable medium of claim 18 wherein said update values, including said update step-size adaptation, are computed according to

$$\Delta_{\iota} \mathbf{W}(\omega) = -\frac{\mu}{h} \nabla_{w} E$$

where  $\mu$  is a fixed learning constant, h is a weighting factor for the step-size adaptation, E is a filter-parameter cost function operating on a square difference respecting a diagonal covariance of said component signals, and E is a gradient step for E.

20. (previously amended) The computer readable medium of claim 15 wherein a running average of said cross correlation estimation values are produced according to

$$\hat{\mathbf{R}}_{x}(t,\omega) = (1-\gamma)\hat{\mathbf{R}}_{x}(t,\omega) + \gamma \mathbf{x}(t,\omega)\mathbf{x}^{H}(t,\omega).$$

where  $x(t,\omega)$  is the mixed signal in the frequency domain.